

Time and Spatial Series and Transforms

- Z- and Fourier transforms
 - Gibbs' phenomenon
 - Convolution
 - Cross- and Auto-correlation
 - Multidimensional transforms
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- Reading:
 - › Sheriff and Geldart, Chapter 15

Z-Transform

- Consider a digitized record of N readings: $U = \{u_0, u_1, u_2, \dots, u_{N-1}\}$. How can we represent this series differently?
- The Z transform simply associates with this time series a *polynomial*:

$$Z(U) = u_0 + u_1z + u_2z^2 + u_3z^3 + \dots$$

- ♦ For example, a 3-sample record of $\{1, 2, 5\}$ is represented by a quadratic polynomial:
 $1 + 2z + 5z^2$.
- In Z-domain, the all-important operation of *convolution* of time series becomes simple multiplication of their Z-transforms:

$$U_1 * U_2 \leftrightarrow Z(U_1)Z(U_2)$$

Fourier Transform

- To describe a polynomial of order $N-1$, it is sufficient to specify its values at N points in Z .
- The *Discrete Fourier transform* is obtained by taking the Z -transform at N points uniformly distributed around a unit circle on the complex plane of z :

$$U(k) = \sum_{m=0}^{N-1} e^{-i\frac{2\pi k}{N}m} u_m, \quad k = 0, 1, \dots, N-1$$

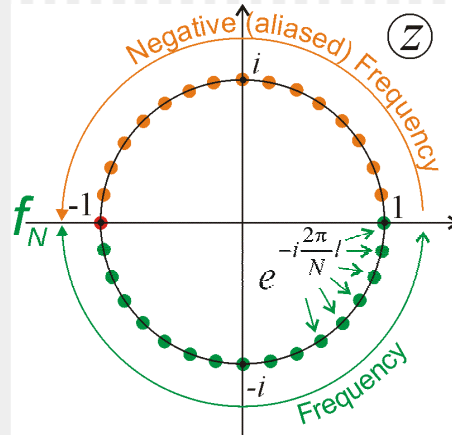
- Each term ($k > 0$) in the sum above is a *periodic function* (a combination of *sin* and *cos*), with a period of N/k sampling intervals:

$$e^{-i\alpha} = \cos\alpha - i\sin\alpha$$

- Thus, the Fourier transform expresses the signal in terms of its frequency components,
 - ♦ and also has the nice property of the Z -transform regarding convolution

Relation of Fourier to Z-transform

- z-points used to construct the Fourier transform:



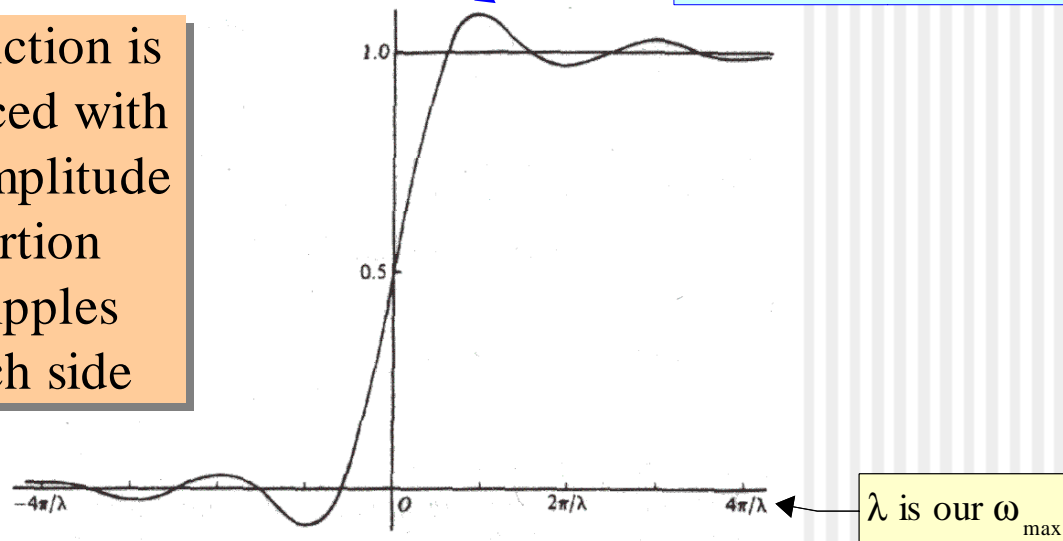
- **Aliasing**: for a real-valued signal, the values of FT at frequencies below and above the Nyquist (orange and green dots) are complex conjugate. Thus, only a half of the frequency band describes the process uniquely.
- This ambiguity is the source of aliasing.
- For this reason, frequencies above f_N should not be used.
- **Note**: forward and inverse FT result in a signal whose N samples are repeated periodically in time.

Gibbs' phenomenon

- At a discontinuity, application of the Fourier forward and inverse transform (with a limited bandwidth), results in ringing.

Note the ~9% “overshoot” at the top and at the bottom

Step function is reproduced with ~18% amplitude distortion and ripples on each side

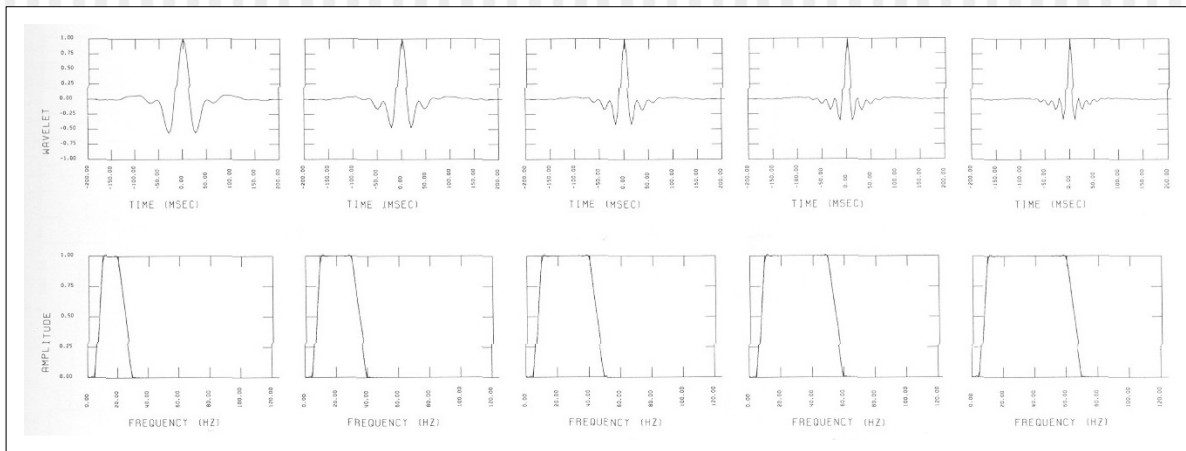


- This is important for constructing time and frequency windows
 - Boxcar windows create ringing at their edges.
 - “Hanning” (cosine) windows are often used to reduce ringing:

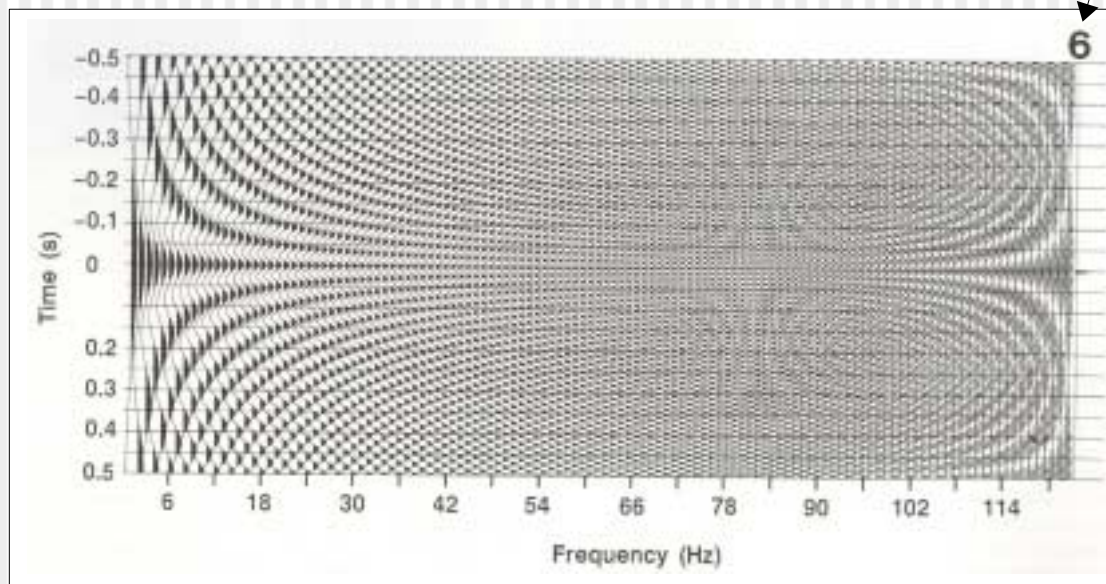
$$H_{\Delta t}(t) = \frac{1}{2} \left(1 - \cos \frac{\pi t}{\Delta t} \right).$$

Spectra of Pulses

- For a pulse of width T s, its spectrum is about $1/T$ Hz in width:



- Equal-amplitude (co)sinusoids from 0 to f_N add up to form a spike:



From Yilmaz, 1987

Integral Fourier Transform

- For continuous time and frequency (infinitesimal sampling interval and infinite recording time), Fourier transform reads:

- Forward:
$$U(\omega) = \frac{1}{\sqrt{2\pi}} \int_{-\infty}^{\infty} dt u(t) e^{-i\omega t}.$$

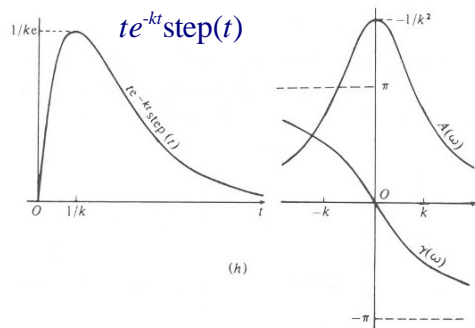
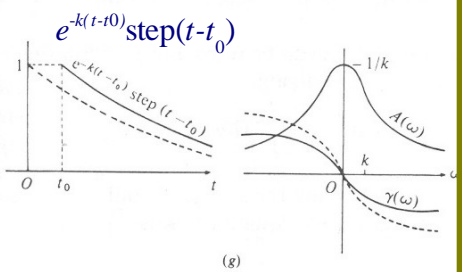
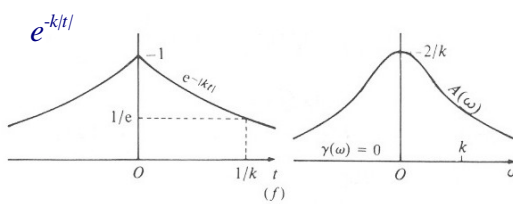
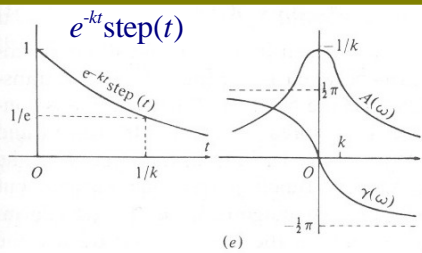
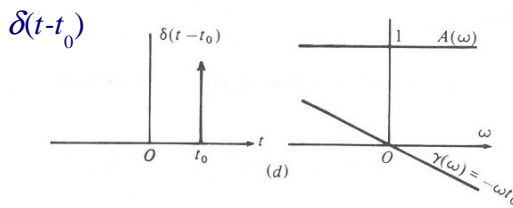
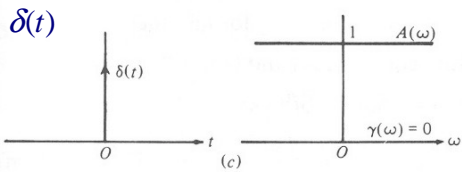
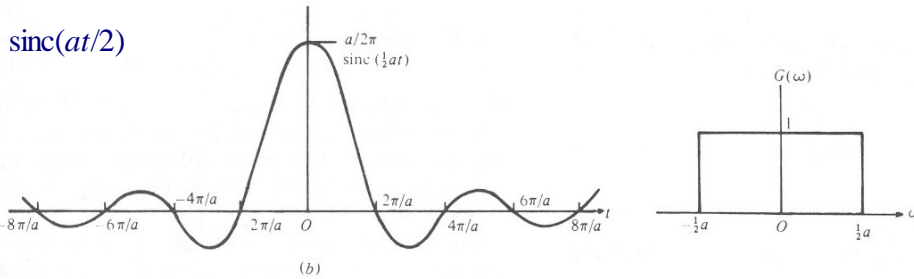
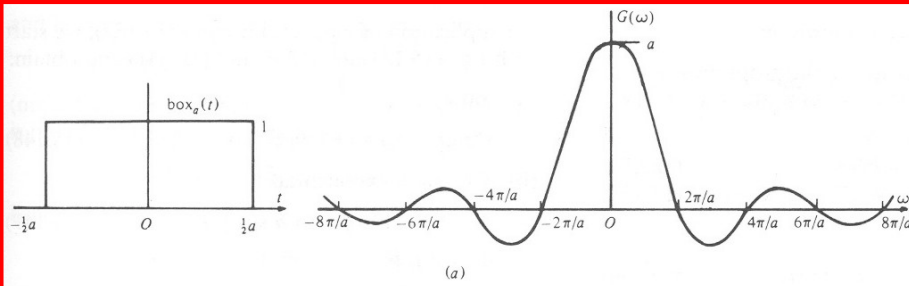
- Inverse:
$$u(t) = \frac{1}{\sqrt{2\pi}} \int_{-\infty}^{\infty} d\omega U(\omega) e^{i\omega t}.$$

- In practice, the bandwidth (and time) is always limited, and so the actual combination of the forward and inverse transforms is rather:

$$u(t) = \frac{1}{2\pi} \int_{-\omega_{max}}^{\omega_{max}} d\omega \left[\int_{-\infty}^{\infty} d\tau u(\tau) e^{-i\omega\tau} \right] e^{i\omega t}.$$

$$u(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} d\tau u(\tau) \left[\int_{-\omega_{max}}^{\omega_{max}} d\omega e^{i\omega(t-\tau)} \right].$$

Sample Fourier Transforms



• Compare the transforms within the boxes...

Convolution

- Convolution of two series, u_i and w_i , denoted $u * w$, is:

$$(u * w)_i = \sum_k u_k w_{i-k}$$

For each i , the result is dot product of u and shifted and “reflected”, or “folded” (i.e., running backwards) w

- In integral form:

$$u(t) * w(t) = \int_{-\infty}^{+\infty} u(\tau) w(t - \tau) d\tau$$

- As multiplication, it is symmetric (commutative):

$$u * w = w * u$$

- Note that to multiply two polynomials, with coefficients u_k and w_k , we would use exactly the first formula above. Therefore, **in Z or frequency domains, convolution becomes simple multiplication** of polynomials (show this!):

$$u * w \leftrightarrow Z(u)Z(w) \leftrightarrow F(u)F(w)$$

- This property facilitates efficient digital filtering.

Cross-Correlation

- Cross-correlation of two series, u_i and w_i , is:

$$(u * w)_i = \sum_k u_k w_{i+k}$$

← Unlike in convolution, no “folding” of w

- In integral form:

$$u(t) * w(t) = \int_{-\infty}^{+\infty} u(\tau) w(t+\tau) d\tau$$

- It is anti-symmetric in the following sense (show this!):

$$(u * w)(t) = (w * u)(-t)$$

- In Z or *frequency* domains, cross-correlation is:

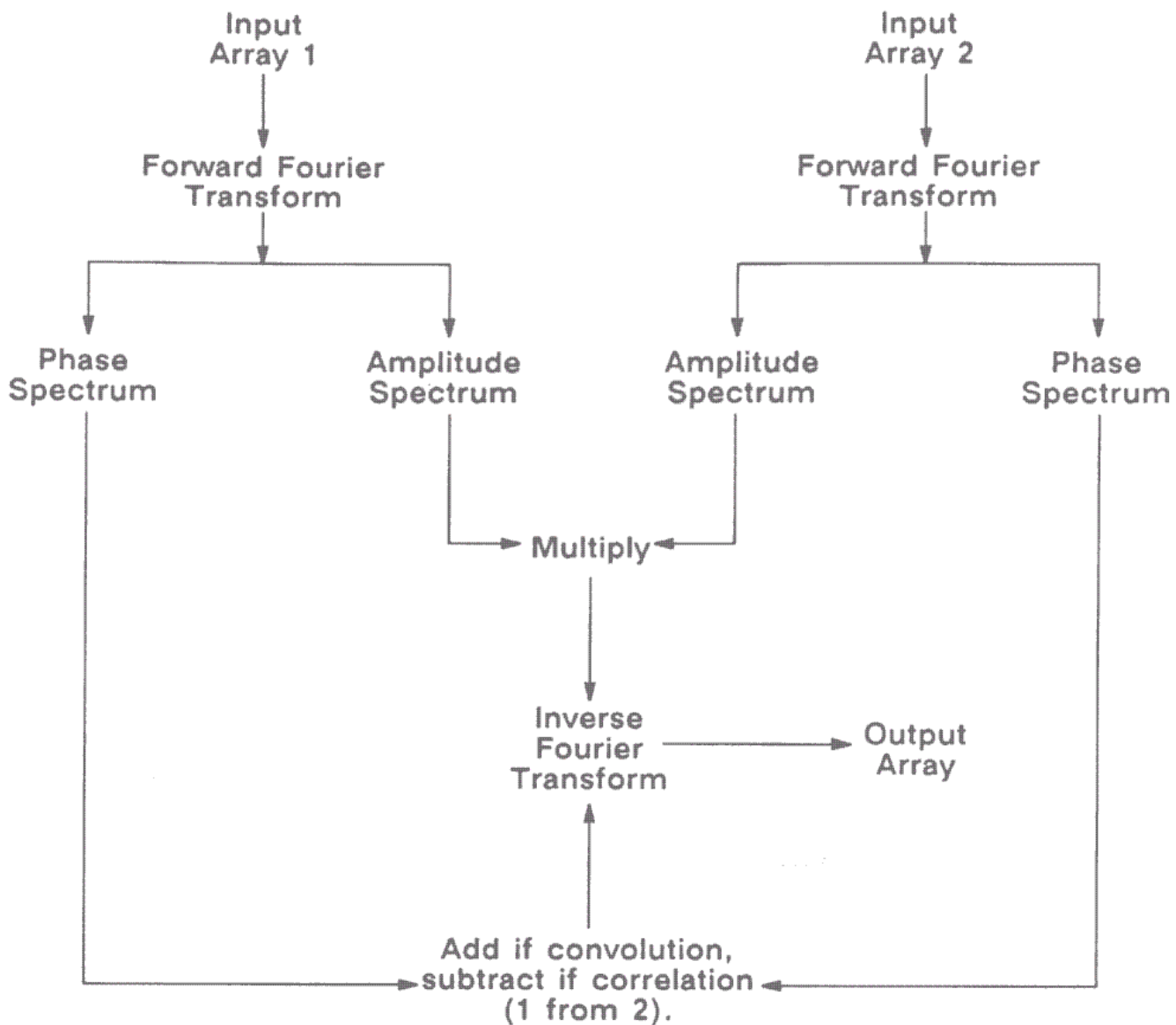
$$(u * w)(z) = \overline{U(z)} W(z)$$

← Complex conjugate

- Cross-correlation is used as a *measure of similarity* between time series.

Practical convolution and cross-correlation

- Is attained via (Fast) Fourier Transform



From Yilmaz, 1987

Autocorrelation

- Cross-correlation of a time function with itself is called *Autocorrelation*:

$$(u * u)_i = \sum_k u_k u_{i+k}$$

- In integral form:

$$Auto_u(t) = \int_{-\infty}^{+\infty} u(\tau) u(t+\tau) d\tau$$

- It always is an even function (show this!):

$$Auto_u(t) = Auto_u(-t)$$

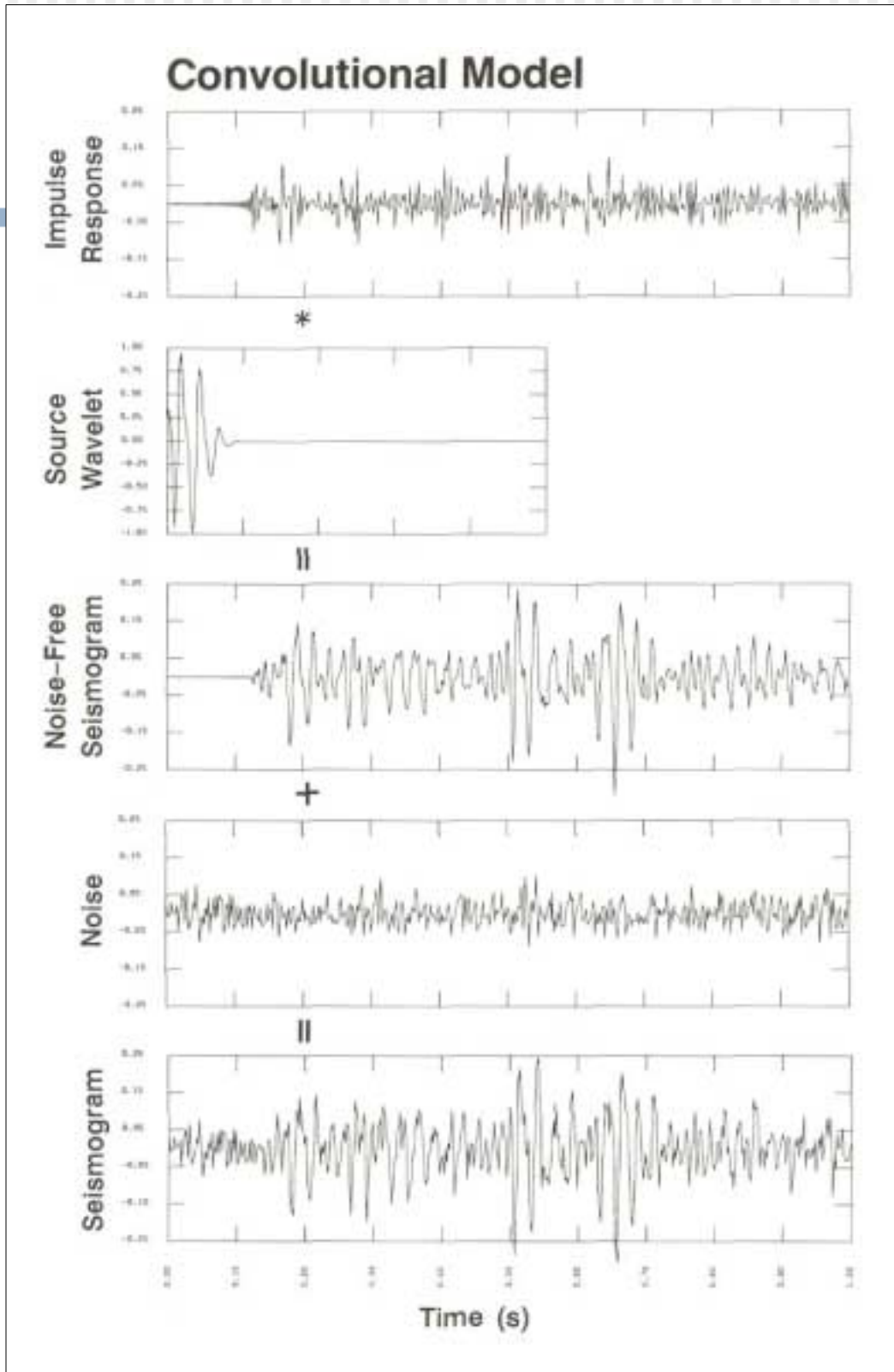
- In Z or *frequency* domains, autocorrelation is:

$$Auto_u(z) = \overline{U(z)} U(z) = |U(z)|^2$$

Always real value -
Energy Spectrum

- Therefore, autocorrelation is also the Fourier transform of the energy spectrum of the signal
 - It is independent of the phase spectrum!
- Autocorrelation is used as a *measure of self-similarity* within a time series.

Convolutional model



From Yilmaz, 1987